

IN THE CLAIMS:

Please amend the claims as follows:

1. (Currently Amended) A method for filtering comprising adaptive filtering an input signal,signal ($x(n)$), interpolating the filtered a filtered signal, interpolating the input signal ($x(n)$) for adapting the adaptive filtering, characterised in thatwherein and adapting the properties of the of an interpolation of the filtered signal,signal are adaptable.
2. (Currently Amended) The method according to claim 1, characterised in thatwherein it comprises comprising providing a reference signal,signal ($d(n)+z(n)$), and combining the combining an interpolated filtered signal and the reference signal for forming an error signal,signal ($e(n)$).
3. (Currently Amended) The method according to claim 2, characterised in thatwherein comprising adapting the properties of the interpolation properties are adapted according to the error signal ($e(n)$) and the interpolated filtered signal,signal ($Y_i(n)$).
4. (Currently Amended) The method according to claim 2 or 3, characterised in thatwherein comprising adapting the properties of the interpolation properties are adapted by changing at least one coefficient of the interpolation.
5. (Currently Amended) The method according to claim 4, characterised in thatwherein comprising adapting the at least one coefficient of the interpolation is adapted by using a normalized least mean square algorithm, wherein the method further comprises using the error signal and the interpolated filtered signal are used as inputs for the algorithm.
6. (Currently Amended) The method according to claim 2any of the claims 2 to 5, characterised in thatwherein it comprising: comprises the following steps:
 - a) computing the filtered output by signal by an equation

$$y(n) = W^t(n)X(n);$$

b) computing the interpolated filtered signal by by an equation

$$Y_I(n) = I^t(n)Y(n);$$

c) adapting the interpolation coefficients of an interpolator by an equation

$$I(n+1) = I(n) + \frac{\mu_I}{\varepsilon + Y^t(n)Y(n)} e(n)Y(n)$$

where μ_I is theis a step-size used to adapt the coefficients of the interpolator, $e(n)$ is theis an output error, $I(n) = [i(n)_1, i(n)_2, \dots, i(n)_M]^t$ is theis an $M \times 1$ vector containing the interpolation coefficients of the interpolator, $Y(n) = [y(n), y(n-1), \dots, y(n - M + 1)]^t$ is theis a vector of the past M samples from the filtered signal $y(n)$, and ε is a small constant;

d) computing the output error $e(n)$ by by an equation

$$e(n) = d(n) + z(n) - y_I(n);$$

e) computing the filtereda filtered input vector $X_I(n)$ by by an equation

$$X_I(n) = \sum_{j=0}^{M-1} i_j X(n-j); \text{ and}$$

f) updating filtering weights by by an equation

$$W(n+1) = F\{W(n) + \mu e(n)X_I(n)\} + q.$$

7. (Currently Amended) The method according to any of the claimsclaim 1 to 6, characterised in thatwherein comprising using finite impulse response filtering is used in said adaptive filtering.

8. (Currently Amended) An apparatus-(1) comprising
an adaptive filter-(2) for filtering an input signal; signal (x(n));,
a first interpolator-(3) for interpolating the filtereda filtered signal; signal,
a second interpolator-(7) for interpolating the input signal, signal (x(n)),
wherein the interpolatedan interpolated input signal is arranged to be used to
adapt the adaptive filter-(2)filter; and characterised in thatwherein the
apparatus-(1) further comprises

a first adapting block-(4) for adapting the properties of the first interpolatorinterpolator-(3).

9. (Currently Amended) The apparatus-(1) according to claim 8, characterised in thatwherein it also comprisescomprising an input-(5-2) for receiving a reference signal.signal-(d(n)+z(n)), and a combiner-(5) for combiningthecombining an interpolated filtered signal and the reference signal for forming an error signal.signal-(e(n)).

10. (Currently Amended) The apparatus-(1) according to claim 9, characterised in thatwherein the interpolation properties are arranged to be adapted according to the error signal-(e(n)) and theand an interpolated filtered signal.signal-(Y_i(n)).

11. (Currently Amended) The apparatus-(1) according to claim 9, claim 9 or 10, characterised in thatwherein the first adapting block-(4) is arranged adapted to change at least one coefficient of the first interpolator-(3)interpolator.

12. (Currently Amended) The apparatus-(1) according to claim 11, characterised in thatwherein the first adapting block-(4) is arranged adapted to use a normalized least mean square algorithm to adapt the at least one coefficient of the first interpolator-(3)interpolator, wherein the error signal and the interpolated filtered signal are arranged to be used as inputs for the algorithm.

13. (Currently Amended) The apparatus-(1) according to claim 8, any of the claims 8 to 11, characterised in thatwherein it also comprisescomprising a second adapting block-(6) for adapting theadapting properties of the adaptive filter.filter-(2).

14. (Currently Amended) The apparatus-(1) according to claim 8,any of the claims 8 to 13, characterised in thatwherein said adaptive filter-(2) is a FIR filter.